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VERIZON CORPORATE SERVICES GROUP INC. C/O CHRISTIAN R. ANDERSEN			AGDEPPA, HECTOR A	
	AN R. ANDERSEN RIDGE DRIVE		ART UNIT	PAPER NUMBER
MAILCODE HQEO3H14			2642	•
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Please find below and/or attached an Office communication concerning this application or proceeding.

		Application No.	Applicant(s)			
Office Action Summary		09/845,486	BERS ET AL.			
		Examiner	Art Unit			
		Hector A. Agdeppa	2642			
The MAILING DATE of this communication appears on the cover sheet with the correspondence address Period for Reply						
A SHO WHIC - Exter after - If NO - Failur Any r	ORTENED STATUTORY PERIOD FOR REPLY CHEVER IS LONGER, FROM THE MAILING DATES as ions of time may be available under the provisions of 37 CFR 1.13 SIX (6) MONTHS from the mailing date of this communication. Period for reply is specified above, the maximum statutory period were to reply within the set or extended period for reply will, by statute, eply received by the Office later than three months after the mailing and patent term adjustment. See 37 CFR 1.704(b).	ATE OF THIS COMMUNICATION 36(a). In no event, however, may a reply be timurily apply and will expire SIX (6) MONTHS from a cause the application to become ABANDONE!	l. ely filed the mailing date of this communication. 0 (35 U.S.C. § 133).			
Status						
2a)⊠	Responsive to communication(s) filed on 19 Au This action is <b>FINAL</b> . 2b) This Since this application is in condition for allower	action is non-final.	secution as to the merits is			
	closed in accordance with the practice under Ex parte Quayle, 1935 C.D. 11, 453 O.G. 213.					
Dispositi	on of Claims	•				
5)□ 6)⊠ 7)□	Claim(s) 1,3,6-10 and 12-35 is/are pending in the 4a) Of the above claim(s) is/are withdraw Claim(s) is/are allowed.  Claim(s) 1, 3, 6-10, 12-35 is/are rejected.  Claim(s) is/are objected to.  Claim(s) are subject to restriction and/or	vn from consideration.				
Applicati	on Papers					
10)	The specification is objected to by the Examiner The drawing(s) filed on is/are: a) acce Applicant may not request that any objection to the of Replacement drawing sheet(s) including the correction The oath or declaration is objected to by the Ex	epted or b) objected to by the Eddrawing(s) be held in abeyance. See ion is required if the drawing(s) is obj	e37 CFR 1.85(a). ected to. See 37 CFR 1.121(d).			
Priority u	nder 35 U.S.C. § 119		•			
12)[/ a)[	Acknowledgment is made of a claim for foreign  All b) Some * c) None of:  1. Certified copies of the priority documents  2. Certified copies of the priority documents  3. Copies of the certified copies of the prior application from the International Bureau ee the attached detailed Office action for a list of	s have been received. s have been received in Application ity documents have been receive I (PCT Rule 17.2(a)).	on No d in this National Stage			
Attachment	(s)					
1) Notice 2) Notice 3) Inform	e of References Cited (PTO-892) e of Draftsperson's Patent Drawing Review (PTO-948) nation Disclosure Statement(s) (PTO-1449 or PTO/SB/08) No(s)/Mail Date	4) Interview Summary ( Paper No(s)/Mail Da 5) Notice of Informal Pa 6) Other:				

Art Unit: 2642

### **DETAILED ACTION**

1. This action is in response to applicant's amendment filed on 8/19/05. Claims 1, 3, 6 - 10, and 12 - 35 are now pending in the present application. **This action is made final.** 

## Claim Rejections - 35 USC § 103

The text of those sections of Title 35, U.S. Code not included in this action can be found in a prior Office action.

This application currently names joint inventors. In considering patentability of the claims under 35 U.S.C. 103(a), the examiner presumes that the subject matter of the various claims was commonly owned at the time any inventions covered therein were made absent any evidence to the contrary. Applicant is advised of the obligation under 37 CFR 1.56 to point out the inventor and invention dates of each claim that was not commonly owned at the time a later invention was made in order for the examiner to consider the applicability of 35 U.S.C. 103(c) and potential 35 U.S.C. 102(e), (f) or (g) prior art under 35 U.S.C. 103(a).

2. Claims 1, 3, 6 – 10, 28 – 30, 34, and 35 are rejected under 35 U.S.C. 103(a) as being unpatentable over US 5,812,638 (Muller) in view of US 6,263,066 (Shtivelman et al.)

Art Unit: 2642

As to claims 1, 3, 6 – 10, 30, and 34, Muller teaches a system and method wherein an incoming call is received at a call center/directory assistance facility and the caller is prompted with a prompting device 24 to provide audio input relating to the purpose for the call. The audio input is recorded by recording device 22 and the recording is provided to an operator/agent for servicing once the recording has been heard by the operator/agent. If no operator/agent is immediately available, the incoming call is put into a queue, and the audio input is stored. Once an operator/agent becomes available, the audio input is provided and the incoming call connected. (Fig. 2, Abstract, Col. 1, lines 49 – 59, Col. 4, line 11 – Col. 5, line 27)

Also, call processing system 10 includes a queuing system 28 which may be included in a switch. (Col. 5, lines 20 - 27) Moreover, any system that can receive an incoming call has a switching device of some sort that actually receives the call. There is no other way to receive a telephony call. The same is true if the claim is read as a caller being at a switching device.

Further note that the prompting device 24 and recording device 22 may be incorporated into an automated directory assistance system (ADAS) comprising a voice response device. Moreover, Muller teaches that a switch may be used to connect an operator/agent to the incoming call in the queue. Therefore, because, as seen in Fig. 2, recording device 22 and prompter 24 are elements separate from the queuing system 28, it is inherent that the incoming call is moved/transferred among those elements.

It is also inherent that some call identifier is associated with the call. Because the recording may be stored, the only way to relay both the queued incoming call and

Art Unit: 2642

the recording to the operator/agent together, at the right time, is to use some identifier to associate the call and the recording.

Finally, Muller teaches that after the communication of incoming call information to the operator, the operator and caller may be connected so that the call may be completed with human-to-human interface. (Col. 8, lines 37 - 44) Therefore, it is clear, that after a call is routed to a voice response device, the call may be transferred back to the switch.

What Muller does not teach assigning a unique call identifier to the incoming call.

However, Muller teaches that well known and already used queuing means may be implemented. Well known queuing methods in the ACD/call center arts include priority queuing/queuing according to agents' skills, etc. as taught by Shtivelman et al. (Col. 9, line 34 - Col. 10, line 37, Col. 10, line 57 - Col. 11, line 19 of Shtivelman et al.) Such well known queuing means queue calls not only in a first-in-first-out basis. Therefore, assigning a unique identifier to the call or using a unique identifier already present in the call such as the ANI or caller ID is inherent. There is no way to queue calls according to priority and skill set among various agents unless this is done. It would have been obvious for one of ordinary skill in the art at the time the invention was made because, as discussed, such queuing means are old and well known and qualify as standard queuing means, which are at least contemplated by Muller. Moreover, ACD/call center calls routinely monitor calls for agent performance, store calls for statistical purposes, etc. For many years now, a caller, before being connected to an agent, will hear an announcement indicating that the call may be recorded for

Art Unit: 2642

various these purposes. If no call identifier were assigned to these calls there would be no way to later retrieve these calls.

As to claim 13, Muller teaches queuing calls with a queuer 28 and routing calls to agents. (Col. 5, lines 9 – 27 of Muller) Shtivelman et al. teaches the same. (Col. 6, lines 8 – 21, 30 – 42, Col. 7, line 38 – Col. 8, line 12, Col. 9, lines 18 – 33 of Shtivelman et al.)

As to claims 14 – 17, see the rejection of claim 1 and also note that it is inherent that some server or some element having server functionality to act as the call processor/router. Even if that element is just the switch discussed above, the switch reads on the claimed generic server with call routing functionality. Muller also teaches that the information requested from a caller is not only the reason for the call such as a directory assistance query, i.e., a telephone number, but also can glean, for example, a language preference and a certain department requested. (Col. 4, lines 35 – 47 of Muller) See also Col. 8, lines 2 – 3 of Shtivelman et al. wherein it is taught that "any" relevant information associated with a caller / call record is retrieved. Finally, Muller teaches that the operator/agent receives the audio input at a device 26 which can perform speech recognition, has a display device 44, among other features. (Col. 5, lines 28 – 67 of Muller) Clearly, without some means for playing the recorded information, an agent would not be able to hear the recorded information. Moreover, a data device is needed since stored audio is data.

As to claims 28 and 29, see the rejection of claim 1 above and note that any ACD operates by forwarding calls to an agent. As described above, Muller teaches accepting

Art Unit: 2642

an incoming call to an ACD, not a particular agent. Of course, there are times when a caller is provided a direct dial number to reach a particular agent, but the standard mode of operation in any ACD is that a caller calls using for example, an 800 number as anyone who has called a call center or directory assistance knows. After being connected to the ACD, the call must be forwarded to an actual agent. In other words, no agent directly answers the call and so connection to an actual agent requires forwarding the call to the agent when on becomes available. See also Col. 5, lines 9 – 14 of Muller wherein Muller teaches "forwarding" the caller information OR communicating it to the operator.

As to claim 35, Muller teaches that the system 10 will alter/manipulate the timing of the audio input recording to match the time it takes the operator/agent to listen/process the recorded information with the time it takes to play a generic/pre-recorded message to the caller so that the caller experiences no dead time in optimal conditions. (Abstract, Col. 5, line 45 – Col. 8, line 36 of Muller) Effectively, the need for an acknowledgement in the system of Muller is erased or interpreted differently, the acknowledgement is integrated into the above-discussed timing feature.

However, the above teachings of Muller are merely a choice to present a user-friendly interface to a caller and such is merely a design choice or preference. It would have been at the least, obvious, to one of ordinary skill in the art at the time the invention was made to have required some acknowledgement that the audio input was heard. The motivation is clear in that there would be no purpose to connecting the

Art Unit: 2642

caller to the operator/agent before they had heard the recorded audio input, or the very purpose of the invention would be defeated.

Remember, that as discussed above, Muller teaches that after communication of incoming caller information to the operator, the operator and incoming caller may actually be connected to effect human-to-human communications. Again, unless the operator has heard the incoming caller's audio input/recorded message there is no point to connecting them. It would completely defeat the purpose of Muller.

3. Claims 18 – 27, and 31 - 33 are rejected under 35 U.S.C. 103(a) as being unpatentable over US 5,812,638 (Muller), in view of US 6,263,066 (Shtivelman et al.), and further in view of US 5,991,390 (Booton).

As to claims 18 - 20, 23, 31, and 32, see the above rejection of claim 1.

Muller and Shtivelman et al. have been discussed above, but what they do not teach is a call center that uses conferencing to connect calls to available operators/agents.

However, many methods of connecting a caller to an operator/agent are known and commonly used methods are that of conferencing. A call center/directory assistance system are also well known as being implemented in a plurality of different environments because the act of connecting a caller to an agent can be thought of as a conference or as being received by the call center. Such a system is taught by Booton. (Col. 10, lines 12 – 26 of Booton)

Art Unit: 2642

It would have been obvious for one of ordinary skill in the art at the time the invention was made to have implemented the call processing system of Muller using conferencing because it is, as discussed, a well known and commonly used method. Moreover, the Muller reference itself, recognizes this and teaches that any number of conventional means may be used to connect the caller to the operator/agent. (Col. 8, lines 42 – 43 of Muller) Such was not discussed in Muller because transferring, sending, conferencing/initiating a separate call, forwarding (as discussed above) are all well known, and all are viable options/methods.

Note that in conferencing, of course to connect the operator/agent to the caller after hearing the recorded audio input is to "bridge" the call.

As to claims 21, 24, and 33, see the rejection of claim 35.

As to claims 22 and 25, as per standard conference calls, the initial connection, which in this case, would be the call at the prompting device 24 and recorder 22, is dropped once the call is bridged between conference participants, i.e., the caller and operator/agent. There is no need for the connection any longer and is a waste of resources besides being the standard method of operation in conference calling.

As to claims 26 and 27, Muller teaches that the incoming call is from the PSTN is inherent or at the least obvious. Muller does not specify that the invention is only to be used in a closed environment such as a PBX. In fact, Muller teaches that any caller can call in and request directory assistance for any city or location which means the callers are at least in some part of the leg, calling from the PSTN. (See the above noted portions of Muller) For this same reason, the voice response device would have to send

Art Unit: 2642

the call over the PSTN. See also, Col. 10, lines 12 – 26 of Booton and note that when an outgoing call is made to a remote agent, again, that call must be made over the PSTN and if any information is collected from the caller using the voice response device of Muller or the interactive voice response (IVR) taught by Booton, as taught by Muller, that information too must be sent over the PSTN to whatever switching device services the remote agent.

See the rejection of claims 18, 19, 23, and 31 and note that as discussed, any of a plurality of well known methods of call connection are contemplated by Muller and especially in the case of call conferencing, it is inherent that the PSTN is used inasmuch as conferencing systems do not merely operate in a closed environment. The purpose of conferencing is to connect parties from disparate networks, locations, etc. The only common network allowing for this connectivity is of course, the PSTN. See also Fig. 1 of Shtivelman et al.

Another motivation for using the PSTN is if remote operators/agents were to be employed. Providing a dedicated connection from the call center/directory assistance center to a remote operator/agent is very costly and many systems merely use the PSTN.

### Response to Arguments

4. Applicant's arguments filed 8/19/05 have been fully considered but they are not persuasive.

Art Unit: 2642

As to applicant's arguments regarding Shitvelman's failure to teach assigning a unique call identifier, examiner's reasoning directed to the Shtivelman teaching more than a standard first-in-first-out basis were directed to applicant's earlier arguments filed 12/28/04. Applicants then asserted that Shtivelman taught no more than first-in-first-out queueing, thereby negating the need to assign any type of identifier to a call as a call could merely be handled in the order it enters the queue. However, as examiner explained, Shtivelman does in fact teach more than standard first-in-first-out queueing because calls may be handled out of order, i.e., placing a new call from a prior customer at the head of the queue or providing the ability for a customer to be called back, record priority bumping, etc. Hence, examiner still maintains that in order for a call to be handled out of order, as to priority overriding the standard first-in-first-out schema, call identifiers MUST be used.

Applicant's arguments merely assert that Shtivelman makes no explicit mention of unique identifiers but never rebuts examiner's contention that unique identifiers are necessary. Applicant furthermore never presents any OTHER manner of handling a call out of order, via priority queuing, etc. without the use of a call identifier. If applicant could present some other way of processing a call in these scenarios, only then perhaps, would examiner's inherency argument be questionable. Moreover, applicant's own prior art submissions, e.g., US 6,108,410 (Reding et al.) US 6,650,748 (Edwards et al.) and US 6,882,641 (Gallick et al.) all teach the use of some type of unique call identifier for queued calls in a call center, thus supporting examiner's assertion on P. 4

Art Unit: 2642

of the previous office action that assigning unique call identifiers for a myriad of purposes is notoriously old and well known in the call center arts.

As to applicant's arguments regarding the audio input, examiner's rejection was made in the previous rejection, and as discussed above, using the argument/interpretation that incoming calls are associated with a unique identifier and that both Muller and Shtivelman teach some form of audio input being associated with an incoming call. Therefore, in fact, the unique identifier becomes associated with the audio input, either directly or indirectly depending on one's interpretation.

As to applicant's arguments regarding lack of motivation to combine Muller and Shtivelman, examiner maintains there is ample motivation to combine. Both Muller and Shtivelman are of the same art, call centers, and even more specifically, call center queuing methods. Both Muller and Shtivelman teach the recording of or input of some audio. It is also well known in the art that call centers may use any of a plethora of queuing methods / means and to use one method taught by Shtivelman in a call center taught by Muller would be more than obvious. Moreover, even if Muller did not teach the use of call identifiers (although again, as examiner argued previously, EVERY call is uniquely identified by its ANI / callerID / etc. and if not, a call could never be processed). Therefore, even if Muller did not "utilize" call identifiers, they are already there and implementing a method that "did utilize" call identifiers would be obvious and would not teach away, nor render the system of Muller inoperable.

As to applicant's arguments regarding an agent hearing the audio input, the ONLY purpose to providing audio input to an agent of a call center is so that the

Art Unit: 2642

agent can now how to service the customer. Even in the abstract of Muller, it states "recording information provided by an incoming caller, replaying the recorded information to the operator to enable the operator to conduct a search..." It would be completely contrary to Muller or any other like system to process a call before an agent has heard the audio input indicating the purpose for the customer's call. Therefore, even though Muller discusses other variations on this idea, i.e., timing an operator's greeting with the recorded input, etc., would be obvious.

Also, applicant must note that obviousness can only be established by combining or modifying the teachings of the prior art to produce the claimed invention where there is some teaching, suggestion, or motivation to do so found either in the references themselves or in the knowledge generally available to one of ordinary skill in the art.

See *In re Fine*, 837 F.2d 1071, 5 USPQ2d 1596 (Fed. Cir. 1988)and *In re Jones*, 958 F.2d 347, 21 USPQ2d 1941 (Fed. Cir. 1992). In this case, because an agent must hear the recorded input before responding to the customer, it would almost be if not actually inherent that the system will wait until the agent has heard the information.

Even in Muller, simply because the invention is couched around creative timing of recordings and greetings to be more convenient for a calling customer, the fact that such timing is done is done so that the operator can hear the customer's recorded information. Therefore, in one interpretation, Muller actually does teach waiting, the indication aspect being incorporated into the timing of the exchange of recorded information and operator greeting. In other words, Muller has found a technique that can predict when an agent would indicate he/she has heard the information. In another

Art Unit: 2642

interpretation, Muller can be seen as preempting the need for an agent to indicate he/she has heard the recorded information, and so, actually requiring the indication was arguably contemplated by Muller and solved and so it would merely take a step backwards to come up with the present invention from Muller.

As to applicant's arguments regarding the switch, see Fig. 1 of Shtivelman, and note that any incoming call and any outgoing call must pass through a switch, either 17 or 21 and so even in a callback, any recorded information or queue information must be sent to the agent who even in calling back a customer must utilize one of the above switches which indicates bi-directional processing. See also Fig. 2 and Col. 9, lines 18 – 34 of Shtivelman and note that once a call is received an processed by an IVR / VRU, it is placed in the queue. A call cannot be made from a queue and must therefore be "transferred" back to the switch in order for the call to be made / completed.

Moreover, as seen with Shtivelman, it is common for call centers to use intelligent peripherals such as IVR / VRU nodes that are separate from the switching element but of course cannot process a call by itself. Therefore, any call to a call center will be received by a switch first. That call will then be sent to the IVR / VRU node or element. Once the IVR / VRU has completed its task(s), i.e., prompting the customer for information and collecting that information, the call must be sent to the queue to await processing. As is known in the art, when that call is to be processed, communications will happen between an agent and the customer over the switch, NOT the queueing element and not the IVR / VRU, hence again, bi-directional communications as claimed and described by the applicant.

Art Unit: 2642

### Conclusion

5. **THIS ACTION IS MADE FINAL.** Applicant is reminded of the extension of time policy as set forth in 37 CFR 1.136(a).

A shortened statutory period for reply to this final action is set to expire THREE MONTHS from the mailing date of this action. In the event a first reply is filed within TWO MONTHS of the mailing date of this final action and the advisory action is not mailed until after the end of the THREE-MONTH shortened statutory period, then the shortened statutory period will expire on the date the advisory action is mailed, and any extension fee pursuant to 37 CFR 1.136(a) will be calculated from the mailing date of the advisory action. In no event, however, will the statutory period for reply expire later than SIX MONTHS from the mailing date of this final action.

6. Any inquiry concerning this communication or earlier communications from the examiner should be directed to Hector A. Agdeppa whose telephone number is 571-272-7480. The examiner can normally be reached on Mon thru Fri 9:30am - 6:00pm.

If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Ahmad F. Matar can be reached on 571-272-7488. The fax phone number for the organization where this application or proceeding is assigned is 571-273-8300.

Art Unit: 2642

Information regarding the status of an application may be obtained from the Patent Application Information Retrieval (PAIR) system. Status information for published applications may be obtained from either Private PAIR or Public PAIR. Status information for unpublished applications is available through Private PAIR only. For more information about the PAIR system, see http://pair-direct.uspto.gov. Should you have questions on access to the Private PAIR system, contact the Electronic Business Center (EBC) at 866-217-9197 (toll-free).

Hector A. Agdeppa Examiner Art Unit 2642

H.A.A. November 10, 2005

> BING Q. BUI PRIMARY EXAMINER